

UNITED STATES PATENT APPLICATION
FOR
AN INTEGRATED BATTERY AND MEDIA DECODER FOR A PORTABLE
HOST DEVICE, AND METHODS OF OPERATING AND MANUFACTURING
THE SAME

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AN INTEGRATED BATTERY AND MEDIA DECODER FOR A PORTABLE
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CROSS REFERENCE TO RELATED APPLICATIONS

This application is a continuation of PCT application PCT/US01/25777, filed August 17, 2001, which claims the benefit of U.S. Provisional Application No. 60/226,459, filed August 17, 2000.

FIELD OF THE INVENTION

The present invention pertains generally to the fields of power supply and media decoding and, more specifically, to a battery pack for a portable host device that includes an integrated audio or video decoder.

BACKGROUND OF THE INVENTION

The popularity of portable devices (e.g., mobile telephones, personal digital assistants (PDAs), notebook computers, cameras etc.) has been fueled by the increased mobility of people within the workplace and the convenience of continual access to information and communications networks (e.g., the Public Switched Telephone Network (PSTN) and the Internet). A large majority of such portable devices rely upon batteries as a power source. Many modern batteries incorporate electronics to monitor the health of the battery, manage charging, etc.

SUMMARY OF THE INVENTION

According to a first aspect of the present invention, there is provided an integrated accessory for a host device. The accessory includes a media decoder, a battery coupled to the media decoder operationally to provide power to the media decoder, and a connector to couple the accessory to a host device. Within the integrated accessory, the battery is coupled to the connector so as to allow the battery operationally to provide power to the host device, in addition to the media decoder.

In one embodiment, the battery, the media decoder and the connector are integrated within a housing that is configured to be removably coupled to the host device.

The host device may be a portable device (e.g., a notebook computer, PDA, a mobile phone, a wristwatch, a camera, etc.).

Other features of the present invention will be apparent from the accompanying drawings and from the detailed description that follows.

BRIEF DESCRIPTION OF THE DRAWINGS

The present invention is illustrated by way of example and not limitation in the figures of the accompanying drawings, in which like references indicate similar elements and in which:

Figure 1 is a block diagram illustrating the first exemplary embodiment of an integrated accessory for a host device, the integrated accessory including both a media decoder and a battery.

Figure 2 is a block diagram illustrating a second exemplary embodiment of an integrated accessory for a host device, the integrated accessory again including both a battery and a media decoder, the media decoder exhibiting a higher degree of integration with peripherals than to the exemplary embodiment illustrated in **Figure 1**.

Figure 3 is a block diagram illustrating further architectural details of a media decoder in the exemplary form of an audio decoder, and more specifically a DSP core, which may be included within any one of the integrated accessories shown in **Figures 1 and 2**.

Figure 4 is a flow chart illustrating a method, according to an exemplary embodiment of the present invention, of operation of an integrated accessory, and provides details regarding interactions between the integrated accessory and a host device.

appreciated that the invention is not limited to such exemplary devices and applications.

Figure 1 is a block diagram illustrating a power-supply accessory 10, according to a first exemplary embodiment of the present invention. The illustrated components of the power-supply accessory are, in one embodiment, integrated within a housing that is configured to be removably coupled to a host device in the exemplary form of a mobile telephone. The power-supply accessory 10 is shown to include a power supply in the form of a battery 12, a media decoder in the exemplary form of an audio decoder 14 and a connector 18 to facilitate the removable coupling of the power-supply accessory 10 to a host device.

The battery 12 may be any one of a number of battery types typically included within battery packs for multiple devices (e.g., a NiCad, NiMh, alkaline or lithium battery). The battery 12 is shown to be coupled to the connector 18 so as to allow the battery operationally to provide power to the mobile telephone. The battery 12 is also shown to be coupled to a power conversion circuit 20 so as to allow the battery operationally to provide power to the audio decoder 14. The power requirements for the mobile telephone and the audio decoder may be different, and the power conversion circuit 20 operates to adjust voltage levels outputted from the battery 12 to a level appropriate to power the audio decoder 14. The output of the power conversion circuit 20 may also be utilized to power the interface and other illustrated peripheral components of the power-supply accessory 10.

The audio decoder 14 is shown to include a media integrated circuit (IC) 22 that in one embodiment incorporates a Digital Signal Processor (DSP) core (discussed in further detail below) and an embedded, non-volatile memory in the form of a FLASH memory 24. In one exemplary embodiment, the media IC 22 may be the MediaStream chip 111, designed by Parthus Technologies PLC of Dublin, Ireland, the chip supporting access of up to 128 MB of NAND flash memory. The FLASH memory 24 stores both a collection of media decompression algorithms in the exemplary form of audio decompression algorithms, as well as compressed media files in the exemplary form of compressed audio files. Examples of such audio decompression (or decoding) algorithms include the MP3 decompression algorithm, 04148.P030

based on the Fraunhofer Institute Implementation, the Advanced Audio Coding (AAC) algorithm based on the Fraunhofer Institute Implementation, the Microsoft Windows Media Decoder, the Qdesign Audio Decoder and the Audible.com audio decoder. The stored and compressed audio files (not shown) may be decompressed or decoded, for example, by any one of the above mentioned decompression (or decoding) algorithms. As the FLASH memory 24 is programmable, the audio decoder 14 may conveniently be upgraded to support a wide range of compression of technologies.

The power-supply accessory 10 is also shown to include removable memory in the exemplary form of a removable FLASH memory card 26. In one embodiment, the media IC 22 supports the MultiMediaCard (MMC) and SmartMedia formats. Other formats that may be supported by the media IC 22 include the SD and Memory Stick formats. The removable FLASH memory card 26 is shown, in the exemplary embodiment, to store audio files.

The media IC 22 executes system software, uploaded from the FLASH memory 24 on boot up, that implements a file system on both the FLASH memory 24 and the removable FLASH memory card 26, whereby audio files are stored in directories (e.g., similar to directories on a personal computer, with which the reader may be familiar).

The audio decompression algorithms stored within the FLASH memory 24 are stored in an area of the memory 24 that is not visible to the user. As will be described in further detail below, under direction of a mobile telephone, the media IC 22 loads an appropriate audio decompression algorithm for a selected audio file from the FLASH memory 24, and awaits further instructions provided via the mobile phone to the media IC 22.

As described above, the audio decoder 14 stores and executes system software (e.g., the MediaStream Platform 1000 system software developed by Parthus Technologies PLC). This system software may implement a master/slave protocol that facilitates data communications between the audio decoder 14 and the host device. More specifically, the data communications may include commands that are provided from the host device to the audio decoder 14 to, for example, control

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operation of the audio decoder 14. The commands may also include parameter set commands to set parameters of the audio decoder 14, and parameter read commands to read parameters of the audio decoder 14. Further examples are provided below. For example, the host device may interrogate the audio decoder 14 for its current status, request information regarding a next audio file type, load an appropriate decoder to decode a specific audio file type, play the audio file, pause playing of the audio file, stop playing of the audio file, skip to the next audio file, etc. Commands may also be provided from the host device to adjust volume and tone, as well adjusting the parameters of any effects algorithms that may be present.

A discussion now follows regarding the connector 18. The connector 18 provides various interfaces between the host device and the power-supply accessory 10. Referring to the exemplary embodiment of the present invention shown in **Figure 1**, the connector 18 provides three interfaces, namely a power interface 25 whereby the battery 12 operationally provides power to the host device, a data interface 27 whereby a digital audio data is outputted from the audio decoder 14 to a digital-to-analog converter (not shown) within the host device, and a control interface 28 via which commands and other instructions are communicated between the host device and the audio decoder 14. The power interface 25 may adjust the power supply to suitable voltage levels for the host device.

Digital audio output from the audio decoder 14 is shown to be provided to both the data interface 27 of the connector 18 for supply to a DAC within the host device, and to a DAC 30 that is included within the power-supply accessory 10. The output of the audio data to the data interface 27 and the DAC 30 is, in one embodiment, via an I²S Bus. The DAC 30 operates to convert the digital audio output from the audio decoder 14 to an analog signal that may be outputted via a jack connector (not shown) to headphones. In one embodiment, the outputs of the DAC 30 is provided to an output amplifier (not shown) that buffers the outputs of the DAC 30 to allow these outputs to drive the headphones.

The exemplary power-supply accessory 10 accordingly provides two options for outputting audio (or other media) to a user. In a first case, the digital audio output from the audio decoder 14 is fed from the power-supply accessory 10 to the

host device via the data interface 27 of the connector 18. This digital audio as received by the host device may then be combined with other digital audio (e.g., telephone call audio) from the host device itself. This allows a single headphone set to be plugged into a jack of the host device. In the exemplary embodiment in which the host device is a mobile telephone, the user can accordingly listen to music and make/receive telephone calls via this single jack. For example, when an incoming call arrives or the user wishes to make a telephone call, the mobile telephone may mute the digital audio output received from the power-supply accessory 10, and route the telephone call audio to the headset. When the call is terminated, or when commanded by the user, the mobile telephone may route audio output received from the power-supply accessory 10 to the headset.

In a second case, the DAC 30 that is integral with the power-supply accessory 10 outputs an analog audio signal that is supplied to a jack connector integral within the power-supply accessory 10 for headphones.

Data communications between the audio decoder 14 and the control interface 28 of the connector 18 are, in the exemplary embodiment, performed via an I²C control bus which is shown in Figure 1 to couple the audio decoder 14 to control interface circuitry 32_a, which enables an external controller (e.g., associated with a headset) to control operation of the audio decoder 14. The power-supply accessory 10 may also optionally include control interface circuitry 32_b through which the audio decoder 14 communicates with the control interface 28 via a SMBUS bus.

The exemplary embodiment of the power-supply accessory 10 shown in Figure 1 also includes a Universal Serial Bus (USB) interface 34, coupled to a USB jack, via which algorithms and songs may be downloaded to, or uploaded from, the audio decoder 14 and the removable FLASH memory card 26.

Audio files may be downloaded to (or uploaded from) the power-supply accessory 10 in a number of ways. Firstly, such audio files may be downloaded (or uploaded) via the host device (e.g., a mobile telephone). For example, it is possible to upload and download compressed audio files from the Internet utilizing a mobile telephone. Although a relatively slow data transfer rates are achievable utilizing current mobile telephones, the next generation of mobile telephones (e.g., G3

Power management is an important consideration for mobile applications. Accordingly, a platform supported by the media IC 22 provides low power consumption (e.g., less than 70mw while playing). In one embodiment, the audio decoder 14 requires a power supply voltage of 1.8 and 3.3V DC, which may be provided by the power conversion circuit 20.

A media platform supported by the media IC 22 may provide a number of power-savings modes that may be entered into under software control to reduce overall power consumption. For example, system software executed by the media IC 22 may implement "wait", "stop" and "power down" states. Each mode removes a clock signal from successively larger portions of the power-supply accessory 10 until, in the "power down" mode, an external crystal amplifier is disabled to completely remove a clock source to the accessory 10. All three modes of operation may be entered into under control of the system software. The "wait" and "stop" modes may be exited on the occurrence of a hardware reset, a debug request, or an unmasked interrupt. The "power down" mode may only be exited by a hardware reset.

Figure 2 is a block diagram illustrating a second exemplary embodiment of the power-supply accessory 10, which differs from the embodiment illustrated in **Figure 1** in that the **Figure 2** embodiment provides a more highly integrated solution. Specifically, a number of the peripheral components of the **Figure 1** embodiment (e.g., the USB interface 34, the DAC 30, the power conversion circuit 20 and the control interface circuitry 32) are integrated on-chip within the media IC 22, and are accordingly not separately illustrated. It will however be appreciated that the function of the **Figure 2** embodiment is substantially similar to the **Figure 1** embodiment.

Figure 3 is a block diagram providing further architectural details regarding the media IC 22, according to an exemplary embodiment of the present invention. Central to the media IC 22 is a DSP core 40 (e.g., the DSP 2410 programmable DSP core designed by Parthus Technologies PLC). Benefits associated with the use of a programmable DSP core 40 (as opposed to a hardware-based architecture) for compressed audio decoding include the use of a programmable memory that

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facilitates the convenient updating of decoding algorithms and control software. For example, DSP program code may be stored within the FLASH memory 24 and uploaded by the DSP core 40 on power-up. This allows for updates as audio decompression standards evolve and for new audio decoding algorithms to be included within the power-supply accessory 10 as these become available. Further, additional effects algorithms (e.g., 3-D surround sound) may conveniently be added.

Various peripherals are provided around the DSP core 40 to implement the audio decoder 14. Specifically, X-RAM, Y-RAM and Program-RAM 42, 44 and 46 support the DSP core 40. A control interface in the exemplary form of an I²C interface 48 facilitates communication with a control unit within a host device (e.g., a mobile telephone). A Serial Peripheral Allow Interface 50 facilitates communications with the removable FLASH memory card 26. A Phase Locked Loop (PLL) provides clock signals for the power-supply accessory 10. A Serial Audio Interface (SAI) 54 is utilized to stream decompressed audio from the media IC 22, in the manner described above, to a data interface 27 of the connector 18, and eventually on to an external DAC incorporated within a host device for conversion to an analog signal to drive headphones.

A FLASH External Memory Interface (EMI) allows the media IC 22 to connect to external memory (e.g., NAND flash and standard SRAM/NOR FLASH memory). This facilitates access to compressed audio files and audio decoder algorithms. Access may also be provided via this interface, for example, to a number of interesting applications, such as applications implementing post-processing effects (e.g., surround sound).

The above-described peripherals allow the DSP core 40 to function as a digital bit-stream compressed audio decoder 14.

In a further embodiment, the media IC 22 may also include a Sony/Phillips Digital Interface Format (SPDIF) interface that allows the power-supply accessory 10 to connect to other devices (e.g., compact disk (CD) players) that support this interface.

With respect to the above-mentioned SAI interface 54, while this interface is most commonly used in output mode, because this interface 54 is under program

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key pay) of the host device. The identifiers for the selected audio files are then communicated, via the control interface 28 of the connector 18, back to the media player (e.g., the audio decoder 14) within the power-supply accessory 10.

At block 70, the media player that examines the selected audio files, and returns file type (e.g., MP3, AAC, WMA, etc.) information identifying compression algorithms whereby the respective audio files have been encoded.

At block 72, the media player then loads appropriate decode algorithms from the FLASH memory 24 for the selected audio files. At block 74, the media player begins decoding of the selected audio files utilizing the loaded decode algorithms, and outputs digital audio to the host device. Referring specifically to **Figure 1**, in this embodiment, the digital audio is outputted from the audio decoder 14 via the data interface 27 of the connector 18. It will also be appreciated that the digital audio may be outputted via the I²S interface to the DAC 30 for direct output from the accessory 10.

At block 76, a DAC (not shown) within the host device converts the digital audio signal into an analog signal, and provides output via a signal reproduction device (e.g., headphones or a speaker) coupled to the host device.

At block 78, the media player provides information embedded within a selected audio file (e.g., song and artist name, etc.) to the host device via the control interface for display to a user.

At block 80, the user may optionally modify parameters of the media player (e.g., the volume, tone, etc. of the digital output of the audio decoder 14) via a user interface provided by the host device.

Figure 5 is a block diagram illustrating an exemplary embodiment of the present invention wherein the power-supply accessory 10 operates as a battery pack, including an integrated media player, for a host device in the exemplary form of a mobile telephone 90. As illustrated, the power-supply accessory 10 includes a housing within which components are integrally housed, and which includes the connector 18 to facilitate removable coupling of the power-supply accessory 10 to the mobile telephone 90. The power-supply accessory 10 is also shown to include a number of plated contacts, coupled to the connector 18. Contact is maintained by 04148.P030

spring pressure (or bias) between contacts 94 (shown in broken line) of the mobile telephone 90 and contacts 92 of the power-supply accessory 10.

The mobile telephone 90 is also shown to include an input interface 96 (e.g., a numeric keyboard, a QWERTY keyboard, a touch pad or the like) and a display interface 98 (e.g., a LCD screen) utilizing which the user can interact with the mobile telephone 90 and the power-supply accessory 10, and be provided with additional information.

The mobile telephone 90 may also include a data input device, in an exemplary form of a microphone 100 or a camera (not shown), and a signal reproduction device in an exemplary form of a speaker 102 or video screen (not shown).

The mobile telephone 90 and the power-supply accessory 10 are each shown to include a jack via which a media signal (e.g., an audio or video signal) may be outputted from the respective component to, for example, a pair of headphones, shown at 106. The power-supply accessory 10 is also shown to include a high-speed data port 107 (e.g., a USB or FireWire jack).

The incorporation of a media player (e.g., the audio decoder 14) within a power-supply accessory 10 (e.g., a battery pack) as illustrated in **Figure 5** is particularly advantageous in that a host device (e.g. the mobile telephone 90) typically includes an input interface (e.g., numeric key pad) and an output interface (e.g., a LCD display or speaker) that can be leveraged to control the media player as integrated within the power-supply accessory 10. Accordingly, costs associated with producing a media player, which leverages existing components in a host device, can be reduced relative to products where such interfaces must be incorporated within the product.

While an audio decoder 14 has been held out as an example of a media player for illustrative purposes in the above exemplary embodiments, it will be appreciated that the media player need not necessarily be an audio decoder (or audio player). Specifically, the media player may include broader functionality, and be capable of decoding (and encoding) both audio and video signals. For example, the media player may operate as both an audio and video encoder and decoder. In these cases,

an appropriate input device of a host device may be utilized to provide input to such a media player, and to reproduce output from such a media player. For example, where a media player within a power-supply accessory 10 is capable of processing video data, a camera (e.g., a digital video camera) included within the host device may be utilized to provide data to the media player for encoding and storage. Similarly, a video display (e.g., a LCD) included within the host device may be utilized to reproduce video signals decoded by, and received from, a media player within the power-supply accessory 10.

For the purposes of illustration, a number of exemplary media player operation commands in the form of MP3 commands that may be provided from a player control application, executing on the host device and provided to the media player within an power-supply accessory 10, are provided in Table 1. Each MP3 status/command variable may be accessed as a single word parameter, and is addressed by an offset supplied by the host device. The MSB of each command is a DSP application number, and in this example, the relevant media player in the form of a MP3 player has been designated an application number of 1.

Table 1

Command Name	Coding(hex) app:cmd:num:arg (MSB:LSB)	Description	Response(hex) app:cmd:stat:num:data (MSB:LSB)
MP3_GET_TRACK_TAG_INFO	0001:00:000003:0000 NN	Returns the track ID of track number #NN = 135 ASCII bytes.	0001:00:SSSSSS:000087:X i,X ₁₃₅ (X ₁ ,X ₁₃₅ = 135 tag character bytes packed into 45 24 bit words)
MP3_GET_PLAY_STATE	0001:01:000000	Returns the play state i.e. 0= PLAYING, 1 = STOPPED, 2 = PAUSED.	0001:01:SSSSSS:000003:X XXXXXX XXXXXX = 0(play- ing),1(stopped),2(paused)
MP3_GET_FILE_STATE	0001:02:000000	Returns TRUE(-1) if the end of file for the current track has been reached.	0001:02:SSSSSS:000003:X XXXXXX XXXXXX = 0x000000 = EOF FALSE XXXXXX = 0xfffff = EOF TRUE
MP3_GET_COMMAND	0001:03:000003:000 OOO= parameter table offset(valid range 1..4410) - see Section .	Returns the value of the internal MP3 parameter with table offset number OOOOOO(an unsigned 24 bit int)	0001:03:SSSSSS:000003:D DDDDDD (DDDDDD = returned data)
MP3_SET_COMMAND	0001:04:000006:000 OOO:DDDDDD	This command allows the host to set the writ- table internal MP3 parameters.	0001:04:SSSSSS:000000

		OOOOOO specifies the internal parameter number (table offset), and DDDDDD is the 24 bit data value to be written	
MP3_PLAY_TRACK	0001:05:000003:NNN NNN	plays track number NNNNNN	0001:05:SSSSSS:000000
MP3_STOP_TRACK	0001:06:000000	stops the currently playing track	0001:06:SSSSSS:000000
MP3_PAUSE_TRACK	0001:07:000000	pauses the currently playing track	0001:07:SSSSSS:000000
MP3_CONTINUE_TRACK	0001:08:000000	continues playing the currently paused track	0001:08:SSSSSS:000000
MP3_OPENFILE_CMD	0001:09:000048:DDD DDD _{0..DDDDDD} ₄₇	sets the filename - see - Section	0001:09:SSSSSS:000000
MP3_FFWD_CMD	0001:0a:000003:DDD DDD	fast forwards/rewinds by DDDDDD bytes - see - Section	0001:0a:SSSSSS:000000

Table 2, below, describes commands that are utilized to set/read internal parameters of a media player in the form of an exemplary MP3 player. All of the MP3 commands listed in **Table 2** are controllable by reading/writing into a shared global parameter area within a DSP address space.

When the DSP core 40, as described above, has finished decoding a block of audio data, it updates operational parameters with a copy of the host parameter area. All command variables are accessed via single 24-bit word values.

For example, to set an internal MP3 decode parameter, the host device sends a MP3_Set_Command with a parameter offset number, followed by a data value to be written. To read an internal MP3 decoder parameter, the host device sends a MP3_Get_COMMAND with the parameter offset, responsive to which a 24-bit parameter is returned.

Table 2

Param Offset ₁₀	Internal MP3 variable(s)	R/W	Description	return data
0	software revision	R	BCD i.e. 0x0100 = version 1.00	24 bit BCD
1	algorithm	R	1 = MPEG layer 1, 2=MPEG layer 2, 3=MPEG layer 3, 4 = MPEG AAC	24 bit int
2	status	R	-1 = "status info not supported", 0 = running 1 = busy (init, sync etc..)	24 bit int

3	error number	R	-1 = "error numbers not supported", 0 = no errors(running)	16 bit int, right justified
4	error counter	R	number of fatal errors since last boot	24 bit unsigned int
5	frame count	R	"sign of life"	24 bit unsigned int
6	set left level	W	0=max vol,1=1.5dB atten,2=3dB atten,3=4.5dB atten etc...	N/A
7	set right level	W	0=max vol,1=1.5dB atten,2=3dB atten,3=4.5dB atten etc...	N/A
8	bit rate	R	nominal overall bitrate	24 bit unsigned int, units bits/sec.
9	PCM sample rate	R	external device sample freq	24 bit unsigned int, units HZ
10	reserved			
11	reserved			
12	mode	R	0=stereo, 1=joint stereo, 2=dual channel, 3=L+R/2, 4=left, 5=right, 6=customised double mono splitting	24 bit unsigned int
13	mode extension	R	0 = no MS and no IS 'joint stereo' 1 = IS only 2 = MS only 3 = MS and IS	24 bit unsigned int
14	emphasis	R	0=none, 1=50/15, 2=CCITT J.17	24 bit unsigned int
15	bitstream protection bit	R	0=none, 1= ISO CRC enabled	24 bit unsigned int
16	bitstream private bit	R	0=set to zero, 1=set to one	24 bit unsigned int
17	bitstream copyright bit	R	0=no copyright, 1=copyright protected	24 bit unsigned int
18	bitstream original bit	R	0=copy,1=original	24 bit unsigned int
19	reserved			
20	reserved			
21	reserved			
22	reserved			
23	reserved			
24	reserved			
25	reserved			
26	reserved			
27	reserved			
28	reserved			
29	reserved			
30	reserved			
31	reserved			
32	reserved			
33	reserved			
34	reserved			
35	Bass enhancement	W	Valid range -12..0.12 Default value = 0dB gain,	N/A
36	Bass frequency	W	Bass enhance frequency,Default value = 250HZ	N/A
37	Treble enhancement	W	Valid range -12..0.12 Default value = 0dB gain,	N/A
38	Treble frequency	W	Treble enhance frequency,Default value = 2500HZ	N/A
39	granule count	R	reset value = 0	24 bit unsigned integer
40	Left channel meter	R	reset value = 0	24 bit unsigned integer
41	Right channel meter	R	reset value = 0	24 bit unsigned integer
42	Frame length in bits	R	reset value = 0	24 bit unsigned integer
43	Bitstream buffer data demand for input buffer	R	reset value = 0x60	24 bit unsigned integer
44	PCM space required in output buffer	R	reset value = 0x60	24 bit unsigned integer

In an exemplary embodiment of the present invention, a command set is also available to a host device to control the media IC 22, which includes the DSP core 40.

Table 3, below, describes a list of exemplary commands that may be available to a host device.

Table 3

Command Name	Coding(hex) app:cmd:num:arg (MSB:LSB)	Description	Response(hex) app:cmd:stat:num:data (MSB:LSB)
DSP_SYS_GET_STATUS	0000:00:000000	requests the DSP system status - returned in SS field in response	0000:00:SSSSSS:000000
DSP_SYS_HOST_STATUS		not supported	
DSP_SYS_GET_SW_ID	0000:01:000000	requests the DSP software revision - see Section	0000:01:SSSSSS:000003:00RRRR (RRRR = BCD revision data)
DSP_SYS_STOP	0000:02:000000	puts DSP into stop mode - can only recover via IRQA or RESET.	no reply expected
DSP_SYS_WAIT	0000:03:000000	Puts DSP into WAIT mode - can recover via any host command.	no reply expected
DSP_SYS_WRITE_MEM		no longer supported - see new functions DSP_WRITE_X_MEM DSP_WRITE_Y_MEM DSP_WRITE_P_MEM	
DSP_SYS_READ_MEM		no longer supported - see new function DSP_READ_X_MEM DSP_READ_Y_MEM DSP_READ_P_MEM	
DSP_SYS_WRITE_REG		not supported	
DSP_SYS_READ_REG		not supported	
DSP_CMD_IF_RESET		not supported	
DSP_TERMINATE_AND_UNLOAD_APP	0000:04:000003:XXXXXX (XXXXXX=app number a 24 bit unsigned int)	This command causes the DSP application with application number XXXXXX to terminate.	0000:04:SSSSSS:000000
DSP_LOAD_AND_LAUNCH_APP	0000:05:000003:XXXXXX (XXXXXX=app number a 24 bit unsigned int)	This command invokes the DSP app loader function. The DSP uses the app number to reference the application code start address in flash.	0000:05:SSSSSS:000000
DSP_READ_NUM_MMC_TRACKS	0000:06:000000	returns the number of tracks on the MMC	0000:06:SSSSSS:000003:nnnnnn (nnnnnn =24 bit unsigned int)
DSP_READ_NUM_FLASH_TRACKS	0000:07:000000	returns the number of tracks in flash	0000:07:SSSSSS:000003:nnnnnn (nnnnnn =24 bit unsigned int)
DSP_GET_AUDIO_FORMAT_TYPE	0000:08:000000	returns the current audio format	0000:08:SSSSSS:000003:fffff (fffff = format type = unsigned 24 bit int) - see Section
DSP_WRITE_X_MEM	0000:09:000006:AAAAAA:	DSP does:	0000:09:SSSSSS:000000

	DDDD:DD AAAAAA = 24 bit address DDDDDD = 24 bit int data	move DDDDDD,a move #AAAAAA,r0 move a,x:(r0)	
DSP_WRITE_Y_MEM	0000:0A:000003:AAAAAA: DDDD:DD AAAAAA = 24 bit address DDDDDD = 24 bit int data	DSP does: move DDDDDD,a move #AAAAAA,r0 move a,y:(r0)	0000:0A:SSSSSS:000000
DSP_WRITE_P_MEM	0000:0B:000003:AAAAAA: DDDD:DD AAAAAA = 24 bit address DDDDDD = 24 bit int data	DSP does: move DDDDDD,a move #AAAAAA,r0 move a,p:(r0)	0000:0B:SSSSSS:000000
DSP_READ_X_MEM	0000:0C:000003:AAAAAA	DSP does: move #AAAAAA,r0 move x:(r0),a return a	0000:0C:SSSSSS:000003:DDDDDD (DDDDDD = 24 bit returned data word)
DSP_READ_Y_MEM	0000:0D:000003:AAAAAA	DSP does: move #AAAAAA,r0 move y:(r0),a return a	0000:0D:SSSSSS:000003:DDDDDD (DDDDDD = 24 bit returned data word)
DSP_READ_P_MEM	0000:0E:000003:AAAAAA	DSP does: move #AAAAAA,r0 move p:(r0),a return a	0000:0E:SSSSSS:000003:DDDDDD (DDDDDD = 24 bit returned data word)
DSP_WRITE_GPIO	0000:0F:000003:00GGDD	writes DD to GPIO number (GG>>8). N.B. only the LSB of the DD byte is actually used e.g. 0000:17:000003:00c388 writes a logic 0 to GPIO pin 3	0000:0F:SSSSSS:000000
DSP_READ_GPIO	0000:10:000003:0000GG	reads GPIO pin GG e.g. 0000:18:000003:000002 returns GPIO pin 2's value	0000:10:SSSSSS:000003:DDDDDD (DDDDDD = 24 bits returned but only the LSB is significant bits 1..23 are set to zero.
DSP_CONFIG_GPIO	0000:11:000003:00GGDD	configures GPIO pin GG as input or output. The LSB of the DD byte is used as :- LSB = 0 = output, LSB = 1 = input e.g. 0000:19:000003:000401 configures GPIO pin 4 as an input (N.B. only GPIO pins 0,1,2 and 8 are controllable from the host) all other values will be ignored.	0000:11:SSSSSS:000000
DSP_READ_MMC_CID_REG	0000:12:000000	returns the 128 bit MMC CID register data right justified (i.e. left most 16 bits are zero padded)	0000:12:SSSSSS:0000012:0000nn: nnnnnn:nnnnnn:nnnnnn:nnnnnn:nnn nnn _{ab}
DSP_READ_MMC_CSD_REG	0000:13:000000	returns the 128 bit MMC CSD register data right justified (i.e. left most 16 bits are zero padded)	0000:13:SSSSSS:0000012:0000nn: nnnnnn:nnnnnn:nnnnnn:nnnnnn:nnn nnn _{ab}
DSP_GET_TOTAL_MMC_MEM_SIZE	0000:14:000000	returns total MMC memory size in bytes	0000:14:SSSSSS:000006:DDDDDD :DDDDDD (DDDDDDDDDDDD = 48 bit unsigned int)
DSP_GET_FREE_MMC_MEM	0000:15:000000	returns total MMC free memory in bytes	0000:15:SSSSSS:000006:DDDDDD :DDDDDD (DDDDDDDDDDDD = 48 bit

DSP MMC_RESET	0000:16:000000	resets the MMC card - pass/fail result is returned in system status word	0000:16:SSSSSS:000000
DSP_READ_MMC_BLOCK	0000:17:000003:NNNNNN NNNNNN = blk num = 24 bit unsigned int)	reads a block of 512 bytes from the MMC	0000:17:SSSSSS:000020:word ₁₇₁ (bytes are packed 3 per 24 bit word - the 2 left most bytes are zero padded)
DSP_WRITE_MMC_BLOCK	0000:18:000024:BBBBBB word ₁₇₁ (= the right most byte of word 171 is zero padded)	writes a block of 512 bytes to MMC block BBBBBB	0000:18:SSSSSS:000000
DSP MMC_ERASE_BLOCKS	0000:19:000006:SSSSSS EEEEEE = start block number, EEEEE = end block number, both 24 bit unsigned quantities)	erases block command	0000:19:SSSSSS:000000
DSP_MUTE	0000:1A:000000	mute track	0000:1A:SSSSSS:000000
DSP_UNMUTE	0000:1B:000000	unmute track	0000:1B:SSSSSS:000000
DSP_GET_APPS_INFO	0000:1C:000000	returns information about which apps are loaded.	0000:1C:SSSSSS:000009:XXXXXX YYYYYY:ZZZZZ XXXXXX = app control word, bit0 (LSB) = system bit app, bit 1 = MP3 etc... 1 = loaded in RAM, 0 = not loaded. YYYYYY and ZZZZZ words used for future expansion)
DSP_GET_TRACK_INFO	0000:1D:000003:NNNNNN (NNNNNN = track number, treated as a 24 bit unsigned int)	returns information about the track number - see Section	0000:1D:SSSSSS:DDDDDD _{0..} DDDDDD _N
DSP_DEBUG_GETDIR	0000:1E:000003:MMMMM M	see - Section	0000:1E:SSSSSS:000006:DDDD Do 1'
DSP_DEBUG_GETFILE	0000:1F:000006:MMMMM M:IIIIII	see - Section	0000:1F:SSSSSS:000007:DDDDDD 0..6
DSP_DEBUG_GETBUF	0000:20:000003:IIIIII	see - Section	0000:20:000206:SSSSSS:EEEEEE: NNNNNN:DDDDDD _{0..} NNNNNN
DSP_DEBUG_SETDIR	0000:21:000003:MMMMM M:IIIIII	see - Section	0000:21:SSSSSS:000000:
DSP_DEBUG_GOTO_PARENT_DIR	0000:22:000003:MMMMM M (MMMMMM = media type)	moves up a directory from the current directory	0000:22:SSSSSS:000000
DSP_DEBUG_GOTO_ROOT_DIR	0000:23:000003:MMMMM M (MMMMMM = media type)	moves to root dir of the specified media	0000:23:SSSSSS:000000
Used by SSL only for debug			

Table 4, below, describes the field information conveyed in a 24-bit DSP status response to commands listed above in Table 3.

Table 4

bit	description	meaning
0(LSB)	DSP ready	0 = not ready, 1 = ready to received commands from host
1,2	response to last command	b2, b1 0 0 last command executed OK

		0 1 error occurred executing last command 1 0 last command was not executed 1 1 reserved
3,4,5,6,7	self diagnostic results	reserved
8..23(MSB)	undefined	undefined

The above “read DSP audio format” command returns audio format information for a current track, according to Table 5, below.

Table 5

Format No.	Format
0	unknown format
1	PCM
2	MP3
3	Advanced Audio Coding(AAC)
4	Windows Media Audio(WMA)
5	Real Audio
6	Dolby Digital AC-3
7	DTS
8	DVD-Audio(MLP)
9	QDesign

A “get track info” command supplies a 24-bit integer (the track number) to the DSP core 40, responsive to which the DSP core 40 reads a play list file, and returns the information set out below in Table 6.

Table 6

data returned	data type	max num data elements
track duration	null terminated string	8 packed chars
song info(artist+title)	null terminated string	129 packed chars
filename	null terminated string	12 packed chars(8.3 format)
associated application number	unsigned 24 bit integer	1 unsigned int

Note also that embodiments of the present description may be implemented not only within a physical circuit (e.g., on semiconductor chip) but also within machine-readable media. For example, the circuits and designs discussed above may be stored upon and/or embedded within machine-readable media associated with a design tool used for designing semiconductor devices. Examples include a net list

